

WHAT IS CLAIMED IS:

1. A signal processing device which decodes a data stream
which includes a first audio data and a second audio data
5 sampled at different respective sampling frequencies of fs_1 and
 fs_2 , where $fs_1 < fs_2$, comprising:

a decoder which is inputted said data stream and separates
said data stream into said first audio data and said second
audio data;

10 a filter which, among said first and second audio data
outputted from said decoder, performs re-sampling upon said
first audio data at the same sampling frequency fs_2 as that of
said second audio data, and suppresses aliasing distortion due
to said re-sampling; and

15 a delay unit which, among said first and second audio data
outputted from said decoder, delays said second audio data by a
delay period equal to a processing period due to said filter.

2. A signal processing device according to claim 1,
20 wherein said decoder separates said data stream, processing unit
thereof corresponding to said processing period in said filter,
into said first and second audio data having original sampling
frequencies, respectively.

3. A signal processing device according to claim 1,
wherein signal processing delay time in said filter corresponds
to a predetermined processing unit of inputted audio data.

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4. A signal processing device according to claim 1,
wherein said filter comprises:

0 a re-sampling circuit which, among the first and second
100 audio data which are outputted from said decoder, performs re-
sampling upon said first audio data having said sampling
0 frequency of fs_1 at said sampling frequency fs_2 as that of said
30 second audio data; and

35 an FIR filter which suppresses aliasing distortion in said
40 first *of* audio data.

45 5. A signal processing device according to claim 1,
50 wherein said second stream of audio data includes at least audio
55 data for a forward right channel and audio data for a forward
60 left channel.

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70 6. A signal processing device according to claim 2,
75 wherein said second stream of audio data includes at least audio
80 data for a forward right channel and audio data for a forward

left channel.

7. A signal processing device according to claim 3,
wherein said second stream of audio data includes at least audio
5 data for a forward right channel and audio data for a forward
left channel.

8. A signal processing device according to claim 1,
wherein said sampling frequency fs_1 is one of 48 kHz and 44.1
10 kHz, and said sampling frequency fs_2 is twice as high as said
sampling frequency fs_1 .

9. A signal processing device according to claim 2,
wherein said sampling frequency fs_1 is one of 48 kHz and 44.1
15 kHz, and said sampling frequency fs_2 is twice as high as said
sampling frequency fs_1 .

10. A signal processing device according to claim 3,
wherein said sampling frequency fs_1 is one of 48 kHz and 44.1
20 kHz, and said sampling frequency fs_2 is twice as high as said
sampling frequency fs_1 .

11. A signal processing device according to claim 1,

wherein:

 said second stream of audio data includes at least audio data for a forward right channel and audio data for a forward 5 left channel;

 said sampling frequency fs_1 is one of 48 kHz and 44.1 kHz; and

 said sampling frequency fs_2 is twice as high as said sampling frequency fs_1 .

12. A signal processing method which decodes a data stream which includes a first audio data and a second audio data sampled at different respective sampling frequencies of fs_1 and fs_2 , where $fs_1 < fs_2$, comprising:

 a decoding step of inputting said data stream and separating said data stream into said first audio data and said second audio data;

 a filtering step of, among said first and second audio data outputted from said decoding step, performing re-sampling upon 20 said first audio data at the same sampling frequency fs_2 as that of said second audio data, and suppressing aliasing distortion due to said re-sampling; and

 a delay processing step of, among said first and second

audio data outputted from said decoder, delaying said second audio data by a delay period equal to a processing period due to said filtering step.

5 13. A signal processing method according to claim 12, wherein said decoding step separates said data stream, processing unit thereof corresponding to said processing period in said filter step, into said first and second audio data having original sampling frequencies, respectively.

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135 14. A signal processing method according to claim 12, wherein said processing period in said filtering step corresponds to a predetermined processing unit of inputted audio data.

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145 15. A signal processing method according to claim 12, wherein said filtering step comprises:

150 a re-sampling step of, among the first and second audio data outputted from said decoding step, performing re-sampling 20 upon the first audio data having said sampling frequency of fs_1 at the said sampling frequency fs_2 as that of the second audio data; and

155 a filtering step of suppressing aliasing distortion in

10 said first audio data.

15 16. A signal processing method according to claim 12,
wherein said second stream of audio data includes at least audio
data for a forward right channel and audio data for a forward
left channel.

20 17. A signal processing method according to claim 12,
wherein said sampling frequency fs_1 is one of 48 kHz and 44.1
kHz, and said sampling frequency fs_2 is twice said sampling
frequency fs_1 .

25 18. A signal processing method according to claim 12,
wherein:

30 said second audio data includes at least audio data for a
forward right channel and audio data for a forward left channel;

35 said sampling frequency fs_1 is one of 48 kHz and 44.1 kHz;
and

40 said sampling frequency fs_2 is twice as high as said
sampling frequency fs_1 .

45 19. An optical disk reproducing device which reproduces
multi-channel audio signals using a signal processing device

according to claim 8, when reproducing an optical disk upon which said first and second audio data, which have been sampled at respective different sampling frequencies fs_1 and fs_2 with $fs_1 < fs_2$, have been recorded as a single stream of audio data.